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d his
     (FILE 'USPAT' ENTERED AT 18:05:14 ON 08 JUL 94)
             92 S SUBBAND (1A) COD?
L1
L2
              2 S BLOCK LENGTH(1A) DECI?
            854 S BLOCK LENGTH OR L2
L3
             37 S ADAPTIVE BIT#(1A)ALLOCAT?
L4
             24 S ALLOW? NOISE LEVEL
L5
         112984 S INDEX
L6
L7
              8 S L1 AND L3
              6 S L4 AND L7
L8
              1 S L5 AND L8
L9
              0 S L6 AND L9
L10
             28 S MINIMUM AUDIBLE
L11
              1 S L8 AND L11
L12
L13
              0 S L12 AND L6
=> d l12 cls,kwic
5,268,685 [IMAGE AVAILABLE]
                                2 CLASSIFICATIONS
                                                         L12: 1 of 1
          341/76
                           OR
     1.
                           XR
     2.
          381/30
                                                         L12: 1 of 1
               5,268,685 [IMAGE AVAILABLE]
US PAT NO:
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ABSTRACT:

A . . . to the determined transient states, or any similar meth od,

alters the quantizing bit numbers allocated to each block by an **adaptive** **bit** **allocation** circuit. The allocated bit numbers

are altered depending upon the transient state of each block to improve

the signal-to-noise ratio. .

SUMMARY:

BSUM(12)

This . . . signal into signal components in plural frequency ranges,

orthogonally transforms the signal components in the respective frequency

ranges and applies **adaptive** **bit** **allocation** to them to quantize them. The apparatus is capable of effectively reducing quantizing noise for which no effective masking is. . .

DRAWING DESC:

DRWD (7)

FIG. 6 shows a **minimum** **audible** level curve synthesized wi

th a masking spectrum.

DETDESC:

DETD(3)

In . . . human sense of hearing. This will be described in det ail below. This technique uses a combination of the technologies of **subband** division **coding** (SBC), adaptive transform coding (ATC), and **adaptive** **bit** **allocation** (APC-AC).

DETDESC:

DETD(6)

A . . . shown in FIG. 2, the bandwidth of the signal component s is broadened and the time resolution is increased (i.e., the **block* * **length** is reduced) with increasing frequency of the frequency range.

In the low frequency range of 0 Hz to 5 kHz,. . .

DETDESC:

DETD(9)

The . . . FFT processing by the FFT circuits 13, 14 and 15 is grouped into critical bands, and is sent to the **adaptive** **bit** **allocation** circuit 18. The audio frequency range is divided in to critical bands to take account of the frequency resolution charact eristic of . . .

DETDESC:

DETD(10)

The . . . The spectral data in the critical band is requantize d using the bit number allocated to the critical band by the **adaptive** **bit**
allocation circuit 18. The quantized spectral data is fed to the output terminal 19.

DETDESC:

DETD(11)

The . . . transient detector output. This alters, by increasing or decreasing, the allocated bit numbers allocated to each critical be and by the **adaptive** **bit** **allocation** circuit 18.

DETDESC:

DETD(36)

The . . . between the energy of each band and the allowed nois e level. The allocated bit number information is sent to the **adapt ive**
bit **allocation** circuit 18, which quantizes the spectral da ta generated by the FFT circuits 13, 14 and 15 using the bit number.

DETDESC:

DETD(37)

The **adaptive** **bit** **allocation** circuit 18 quantizes the spectral data in each critical band using bit numbers allocated depending on the difference between the. . .

DETDESC:

DETD(38)

The synthesis circuit 27 synthesizes data indicating the so-calle d
minimum **audible** level curve RC and the masking spectrum MS, as
shown in FIG. 6. The **minimum** **audible** level curve is anothe r
characteristic of the human sense of hearing and is generated by t
he
minimum **audible** level curve generator 32. If the absolute noise
level is below the **minimum** **audible** level curve, this noise cannot
be heard. For a given quantization, the **minimum** **audible** le
vel
curve varies, depending on the loudness level when the signal is

reproduced. However, since the manner in which music. . . band in the vicinity of 4 kHz, it can be assumed that quantization noise less the level of the **minimum** **audible** curve will be inaudible i n other frequency bands. Accordingly, when the noise level in the vicinity kHz corresponding. . . the word length set by the system is not heard, the allowed noise level can be provided by synthesizing the **mini mum** **audible** curve RC and the masking spectrum MS. The resulting al noise level in each critical band may be up to the level indicated by the cross-hatched portion in FIG. 6. In this embodiment, the level of **minimum** **audible** curve at 4 kHz is set to correspond to the minimum level corresponding to quantizing using, e.g., 20 bits. FI G.. .

DETDESC:

DETD(40)

The . . . correction is made is as follows: there are instance s where the total number of bits allocated by applying a temporary **adapt ive**
bit **allocation** to all blocks to which bits are to be alloc ated
may be in error with respect to the number of . . .

DETDESC:

DETD(43)

The . . . at the same intensity as that of a pure sound at 1 k Hz. The equi-loudness curve is similar to the **minimum** **audible** leve 1 curve RC shown in FIG. 6. In the equi-loudness curve, for example, a sound in the vicinity of 4. . .

DETDESC:

DETD(44)

It . . . processing apparatus for compressing a digital speech signal

or a digital video signal, etc Further, the synthesis processing for the

minimum **audible** level curve may be omitted. In this case, **minimum** **audible** level curve generator 32 and synthesis cir cuit 27

are unnecessary, and the output of the subtractor 24 is deconvolut ed by.

CLAIMS:

CLMS(3)

3. The apparatus of claim 1, wherein:

each block has a **block** **length**; and the time dividing means divides the signal components in time int

blocks so that block lengths decrease with increasing frequency.

=> d 19 cls,kwic

5,268,685 [IMAGE AVAILABLE] 2 CLASSIFICATIONS L9: 1 of 1

OR

1. 341/76

2. 381/30 XR

US PAT NO: 5,268,685 [IMAGE AVAILABLE] L9: 1 of 1

ABSTRACT:

A . . . frequency bands, and are also sent to a transient detector,

which determines a transient state for each blocks. Changing an **allowed** **noise** **level** calculating circuit in response to the

determined transient states, or any similar method, alters the quantizing

bit numbers allocated to each block by an **adaptive** **bit**
allocation circuit. The allocated bit numbers are altered depending

upon the transient state of each block to improve the signal-to-no ise

ratio. . .

SUMMARY:

BSUM(12)

This . . . signal into signal components in plural frequency ranges,

orthogonally transforms the signal components in the respective frequency ranges and applies **adaptive** **bit** **allocation** to them to quantize them. The apparatus is capable of effectively reducing quantizing noise for which no effective masking is. . .

DRAWING DESC:

DRWD(4)

FIG. 3 is a block diagram showing an example of the **allowed**
noise **level** calculating circuit 20 in the apparatus shows
in FIG.
1.

DETDESC:

DETD(3)

In . . . human sense of hearing. This will be described in det ail below. This technique uses a combination of the technologies of **subband** division **coding** (SBC), adaptive transform coding (ATC), and **adaptive** **bit** **allocation** (APC-AC).

DETDESC:

DETD(6)

A . . . shown in FIG. 2, the bandwidth of the signal component s is broadened and the time resolution is increased (i.e., the **block* * **length** is reduced) with increasing frequency of the frequency range.

In the low frequency range of 0 Hz to 5 kHz,. . .

DETDESC:

DETD(9)

The . . . FFT processing by the FFT circuits 13, 14 and 15 is grouped into critical bands, and is sent to the **adaptive** **bit** **allocation** circuit 18. The audio frequency range is divided in to critical bands to take account of the frequency resolution charact eristic of . . .

DETDESC:

DETD(10)

The **allowed** **noise** **level** calculating circuit 20 calcul ates an
allowed **noise** **level** for each critical band on the basi s of
the spectral data in the critical band, taking account of masking.
Then,
an allocated bit number is calculated for each critical band on the basis
of the **allowed** **noise** **level** and the energy or the peak
value
in the critical band. The spectral data in the critical band is
requantized using the bit number allocated to the critical band by
the
adaptive **bit** **allocation** circuit 18. The quantized spec
tral
data is fed to the output terminal 19.

DETDESC:

DETD(11)

The **allowed** **noise** **level** calculating circuit 20 is sup plied
with the output of the transient detector 17. The **allowed** **no ise**
level calculating circuit corrects the **allowed** **noise**
level for each critical band in response to the transient dete ctor
output. This alters, by increasing or decreasing, the allocated bi t
numbers allocated to each critical band by the **adaptive** **bit*

* **allocation** circuit 18.

DETDESC:

DETD(24)

FIG. 3 is a block diagram showing the configuration of a practica l example of the **allowed** **noise** **level** calculating circuit 20. In FIG. 3, the input terminal 21 is supplied with the spectral data in the frequency domain. . .

DETDESC:

DETD(29)

The . . . is fed into the subtractor 24. The subtractor 24 det ermines a level .alpha. in the convoluted region corresponding to the **al lowed**
noise **level**. This will be described later. The level .alph a.
corresponding to the **allowed** **noise** **level** is the level that will become the **allowed** **noise** **level** in each critical b and after carrying out the deconvolution processing that will be described below.

DETDESC:

DETD(31)

When . . . the critical band is i, the number of the lowest-fr equency critical band being 1, the level .alpha. corresponding to the **al lowed**
noise **level** is determined by the following equation:

DETDESC:

DETD(36)

The . . . by the allowed noise corrector 30. The corresponds to the level difference between the energy of each band and the **allowed **

noise **level**. The allocated bit number information is sent to the

adaptive **bit** **allocation** circuit 18, which quantizes the spectral data generated by the FFT circuits 13, 14 and 15 using the bit number. . .

DETDESC:

DETD(37)

The **adaptive** **bit** **allocation** circuit 18 quantizes the spectral data in each critical band using bit numbers allocated depending on the difference between the energy of each critical band and the

respective **allowed** **noise** **level**. The delay circuit 29 is provided to delay the bark spectrum SB from the energy calculating circuit 22 to take. . .

DETDESC:

DETD(38)

The . . . level in the vicinity of 4 kHz corresponding to the word length set by the system is not heard, the **allowed** **noise** **level** can be provided by synthesizing the minimum audible curve RC and the masking spectrum MS. The resulting **allowed** **noise** **level** in each critical band may be up to the level indicated by the cross-hatched portion in FIG. 6. In this. . .

DETDESC:

DETD(39)

The **allowed** **noise** **level** corrector 30 corrects the
allowed **noise** **level** at the output of the subtractor 28
in
response to information sent from the correction information output
circuit 33. The. . . state determined by the transient detector
17 for
the signal component in each block shown in FIG. 2. Thus, the **al
lowed**
noise **level** a the output of the subtractor 28 is corrected
so
that the bit allocation for each block is corrected according. .

DETDESC:

DETD(40)

The **allowed** **noise** **level** is corrected in response to information indicating an error between the detected quantity of o utput bits used for quantizing by. . . correction is made is as follows: there are instances where the total number of bits allocated by applying a temporary **adaptive** **bit** **allocation** to all blocks to which bits are to be allocated may be in error with respect to the number

r of.

DETDESC:

DETD(41)

To . . . the correction information output circuit 33 provides , in response to the error, data indicating a correction value to correct the **allowed** **noise** **level** at the output of the subtractor 28 , e.g., in response to mode information indicating the transient state of each block. The correction value described above is transmitted to the allowed noise corrector 30 to correct the **allowed** **noise** **level** at the output of the subtractor 28.

DETDESC:

DETD(42)

The correction information output circuit 33 can also provide correction information based on the so. called equi-loudness curve. The **all owed**
noise **level** at the output of the subtractor 28 is corrected in response to correction information that takes account of modes indicating. . .

DETDESC:

DETD(43)

The . . . kHz sound to be perceived as having at the same inte nsity as the 1 kHz sound. For this reason, the **allowed** **noise** **l evel** should have a frequency characteristic given by a curve corresponding to the equi-loudness curve. It can be seen that correcting the **allowed** **noise** **level** by taking the equi-loudness curve into conside ration adapts the system to the characteristics of the human sense of hea ring.

CLAIMS:

CLMS(3)

3. The apparatus of claim 1, wherein:
each block has a **block** **length**; and
the time dividing means divides the signal components in time int

blocks so that block lengths degrees with ingressing frequency

blocks so that block lengths decrease with increasing frequency.

CLAIMS:

CLMS(4)

4. . . .
critical bands, and
the bit allocating means allocates a quantizing bit number for each
critical band on the basis of an **allowed** **noise** **level**
for
each critical band.

CLAIMS:

CLMS(6)

6. The apparatus of claim 4, wherein the bit number altering mean s alters the bit numbers by changing the **allowed** **noise** **lev el** in response to the transient state determined by the transient state determining means.

CLAIMS:

CLMS(10)

into critical bands;
the bit allocation means allocates a quantizing bit number for ea
ch
 critical band on the basis of an **allowed** **noise** **level**
for
 each critical band; and
bit number altering means alters the bit number by changing the
 allowed **noise** **level** in response to the mode number f
or each

CLAIMS:

block.

Page 11

CLMS (15)

15. . . . into critical bands;

the bit allocation means allocates a quantizing bit number for each

critical band on the basis of an **allowed** **noise** **level** for

each critical band; and

the bit number altering means alters the bit allocation by changing the

allowed **noise** **level** in response to the mode number f or each

block.

=> d 18 cit 1-6\box{3}

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2. 5,268,685, Dec. 7, 1993, Apparatus with transient-dependent bi

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381/30 [IMAGE AVAILABLE]

3. 5,230,038, Jul. 20, 1993, Low bit rate transform coder, decode r, and encoder/decoder for high-quality audio; Louis D. Fielder, et al., 395/2; 381/35, 36, 37 [IMAGE AVAILABLE]

5,222,189, Jun. 22, 1993, Low time-delay transform coder, decoder,
and encoder/decoder for high-quality audio; Loius D. Fielder, 395/2;
381/30; 395/2.38, 2.39 [IMAGE AVAILABLE]

5. 5,142,656, Aug. 25, 1992, Low bit rate transform coder, decode r, and encoder/decoder for high-quality audio; Louis D. Fielder, et al., 381/37

[IMAGE AVAILABLE]

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364/715.04, 748; 375/122; 381/29 [IMAGE AVAILABLE]=> d his
     (FILE 'USPAT' ENTERED AT 18:05:14 ON 08 JUL 94)
L1
             92 S SUBBAND (1A) COD?
L2
              2 S BLOCK LENGTH(1A) DECI?
            854 S BLOCK LENGTH OR L2
L3
             37 S ADAPTIVE BIT#(1A)ALLOCAT?
L4
             24 S ALLOW? NOISE LEVEL
L5
         112984 S INDEX
L6
L7
              8 S L1 AND L3
              6 S L4 AND L7
L8
L9
              1 S L5 AND L8
              0 S L6 AND L9
L10
             28 S MINIMUM AUDIBLE
L11
L12
              1 S L8 AND L11
L13
              0 S L12 AND L6
L14
            499 S BIT(1A) ALLOCAT?
            873 S BIT#(1A)ALLOCAT?
L15
L16
            873 S L14 OR L15
             46 S L11 OR L5
L17
L18
              1 S L8 AND L17
              1 S L7 AND L17
L19
              0 S L6 AND L19
L20
L21
             32 S L6 AND L1
L22
             14 S L16 AND L21
L23
              1 S L22 AND L17
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41/76;
381/30 [IMAGE AVAILABLE]
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atus
with time domain and frequency domain block floating; Kenzo Akagir
i,
341/50; 381/37 [IMAGE AVAILABLE]